## IN THE CLAIMS

Please amend the claims as follows:

- 1. (original) A method for generating training data  $(D_T)$  for an automatic speech recogniser (2) for operating at a particular first sampling frequency  $(f_H)$ , comprising the following steps:
  - deriving spectral characteristics ( $S_L$ ) from audio data ( $D_L$ ) sampled at a second frequency ( $f_L$ ) lower than the first sampling frequency ( $f_H$ );
  - extending the bandwidth of the spectral characteristics  $(S_L) \ \ \text{by retrieving bandwidth extending information } (I_{BE}) \ \ \text{from}$  a codebook (6);
  - processing the bandwidth extended spectral characteristics ( $S_{\text{LE}}$ ) to give the required training data ( $D_{\text{T}}$ ).
- 2. (original) A method according to claim 1, where the conversion of audio data  $(D_H, D_L)$  into sets of spectral characteristics  $(S_H, S_L)$  comprises calculating the FFT of the audio data  $(D_H, D_L)$  to give a set of Fourier coefficients (31) and filtering the output of the FFT with a filterbank (22) to give a set of filterbank power values (32).

- 3. (original) A method according to claim 2, where the conversion of audio data ( $D_H$ ,  $D_L$ ) into sets of spectral characteristics ( $S_H$ ,  $S_L$ ) comprises processing the FFT coefficients (31) or the filterbank power values (32) to give a set of log-spectral coefficients (33).
- 4. (currently amended) A method according to any of claims 1 to 3claim 1, where the processing of bandwidth extended spectral characteristics ( $S_{L,E}$ ) comprises a step of altering the spectrum to adjust signal properties of the audio data ( $D_L$ ).
- 5. (original) A method according to claim 4, where the step of altering the spectrum to adjust the signal properties of the audio data  $(D_L)$  is performed in the linear domain.
- 6. (currently amended) A method according to any of claims 1 to  $\frac{1}{2}$  5 claim 1, where the derivation of the spectral characteristics ( $S_L$ ) from audio data ( $D_L$ ) is followed by a step subtracting the mean spectrum from the spectral characteristics ( $S_L$ ).
- 7. (currently amended) A method for training an automatic speech recognition system (2) wherein the data ( $D_T$ ) used for training are at least partially generated using a method according to  $\frac{any}{c}$

## claims 1 to 6 claim 1.

- 8. (original) A method for generating a codebook (6) for extending the bandwidth of spectral characteristics  $(S_L)$  for audio data  $(D_L)$  sampled at a second sampling frequency  $(f_L)$  to spectral characteristics  $(S_H)$  for a first sampling frequency  $(f_H)$  higher than the second sampling frequency  $(f_L)$ , comprising the following steps for each entry of the codebook (6):
  - deriving a first set of spectral characteristics ( $SC_H$ ) from audio data ( $DC_H$ ) sampled at the first sampling frequency ( $f_H$ );
  - performing a sampling rate transformation on the audio data ( $DC_H$ ) to the second sampling frequency ( $f_L$ ) and deriving a corresponding second set of spectral characteristics ( $SC_L$ );
  - generating a codebook entry (12) based on the second set of spectral characteristics ( $SC_L$ ) and augmenting the codebook entry (12) with additional higher frequency information from the first set of spectral characteristics ( $SC_H$ ).
- 9. (original) A method according to claim 8, where augmenting the codebook entry (12) comprises extracting information from the

corresponding first set of spectral characteristics  $(S_{\tt H})$  pertaining to the frequencies above the second sampling frequency  $(f_{\tt L})$  and attaching this information to the codebook entry (12) in the codebook (6).

- 10. (currently amended) A method according to claim 8—or—9, where the derivation of the second set of spectral characteristics ( $SC_L$ ) is followed by a background noise reduction and/or channel normalization step.
- 11. (original) A method according to claim 10, where the spectral characteristics ( $SC_L$ ) comprise a log-spectral representation, and the channel normalization is performed by subtracting the mean log spectrum from the log spectral characteristics ( $SC_L$ ).
- 12. (original) A method according to claim 10, where the spectral characteristics ( $SC_L$ ) comprise a linear spectral representation, and the background noise reduction is performed by subtracting a background noise spectrum from the linear spectral characteristics ( $SC_L$ ).
- 13. (currently amended) A method according to claim 11—and 12, where the background noise reduction is performed by subtracting a

background noise spectrum from a linear spectral characteristics  $(SC_L)$  subsequently calculating the logarithm and then subtracting the mean log spectrum from the log spectral characteristics

- 14. (original) A system (1) for generating training data ( $D_T$ ) for an automatic speech recogniser (2) operating at a particular first sampling frequency ( $f_H$ ), comprising:
  - a converter (3) for deriving spectral characteristics  $(S_L) \ \text{from audio data } (D_C) \ \text{sampled at a second frequency } (f_L)$  lower than the first sampling frequency  $(f_H)$ ;
  - a retrieval unit (4) for retrieving bandwidth extending information for the spectral characteristics  $(S_L)$  from a codebook (6);
  - a processing module (7) for processing the bandwidth- extended spectral characteristics  $(S_{L,E})$  to give the required training data  $(D_T)$ .
- 15. (original) A system (5) for generating a codebook (6) for extending the bandwidth of a set of spectral characteristics  $(S_L)$  for audio data  $(D_L)$  sampled at a second sampling frequency  $(f_L)$  to a set of spectral characteristics  $(S_H)$  for a first sampling frequency  $(f_H)$  higher than the second sampling frequency  $(f_L)$ , comprising:
  - a converter (9) for deriving a first set of spectral

- characteristics ( $SC_H$ ) from audio data ( $DC_H$ ) sampled at the first sampling frequency ( $f_H$ );
- a module (10) for performing a sampling rate transformation on the audio data (DC<sub>H</sub>) to the second sampling frequency( $f_L$ ) and for deriving a corresponding second set of spectral characteristics (SC<sub>L</sub>) for the second sampling frequency ( $f_L$ );
- a codebook entry generator (11) for generating an entry (12) for the codebook (6) based on a second set of spectral characteristics ( $SC_L$ ) and for augmenting the codebook entry (12) with additional higher frequency information from the corresponding first set of spectral characteristics ( $SC_H$ );